

Preamble

- Deliver sound wave (or electrical signal) to a system
- If the system reproduces the waveform faithfully, the signal is undistorted
- If the shape of the waveform is altered, the signal is distorted

Ch7-1

Three Types of Distortion

- Frequency
- Transient
- Amplitude

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FREQUENCY DISTORTION

- Panel A: Transfer function (amplitude response) of a system is the output when the input has equal amplitude at all frequencies.
- Panel B: No distortion at output of system
- Panels C & D
 - ◆ **NOT** all frequencies reproduced the same
 - ◆ System was frequency selective; filtering occurred
 - ◆ Signal underwent frequency distortion

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FREQUENCY DISTORTION

- The three functions in panels B, C, & D describe the system transfer function, or amplitude response, of the system
- The amplitude response reveals evidence of frequency distortion
- Linear systems (A_{in} A_{out}) produce frequency distortions.

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Effects of Rise -Decay Time on Amplitude Response

- Amplitude rises over time from zero to max, and falls over time from max to zero
- Panel A shows amplitude envelope in the time domain
- Panel B shows amplitude spectrum in the frequency domain (frequency spread)
- Initiation & termination of signal produces transients

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Effects of Rise -Decay Time on Amplitude Response

- ◆ The longer the rise-fall time, the less the transient distortion
- ◆ ANSI S3.6-1989
 - >> Time from -20 dB to -1 dB shall not be less than 20 ms

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TRANSIENT DISTORTION

- **A** sine wave of infinite time.
- **B** sin of finite time. (100 ms)
- **C** sin of short time. (4 ms)
- Energy spread to surrounding frequencies, and amplitude spectrum is continuous
- Why?
 - ◆ Its duration is finite
- Nulls at integer multiples of reciprocal of duration and width of each lobe is inversely proportional to duration
- (1/1=10 and 1/0.04=250)

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AMPLITUDE DISTORTION

- Panel A: Do some of the instantaneous amplitudes of the input signal exceed the limits of linearity of the system?
 - ☑ Yes
- What is the result?
 - ☑ The instantaneous amplitudes at or near maximum amplitude are "clipped off"-- output amplitude is not proportional to input amplitude -- the signal has been peak clipped; distortion
- Panel B: more severe peak clipping; more severe distortion

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AMPLITUDE DISTORTION

- The distortion is called amplitude distortion; Why?
 - ☑ Amplitude is the waveform parameter that was altered
- Amplitude distortion is also called nonlinear distortion; Why?
 - ☑ The distortion arose from operating on the nonlinear portion of the I/O function

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Effects of Amplitude, or Nonlinear, Distortion on Amplitude Spectrum

- Input signals in panels A & B are sinusoids
- Output signals are what kind of sound waves?
 - ☑ Complex periodic waveforms
- At what frequencies should you expect to see energy in a complex periodic waveform?
 - ☑ Harmonics
- Thus, amplitude, or nonlinear, distortion also can be called harmonic distortion (if the input waveform is sinusoidal)

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Effects of Amplitude, Nonlinear, or Harmonic Distortion on Amplitude Spectrum

- The figure displays the output spectrum from a nonlinear system
- At the input, $f = 100$ Hz
- At the output, note energy at harmonics of f_0

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Amplitude Response and Dynamic Range

- Panel A: I / O functions for several driving frequencies from .1 kHz to 10 kHz
 - On each I / O function, location of the filled dot identifies point of maximum permissible harmonic distortion
- Panel B: The dots from I / O functions are connected and redrawn in the frequency domain to form the upper curve
 - That defines the amplitude response of the system
 - Electrical noise floor (ENF) of system is shown by lower curve

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Amplitude Response and Dynamic Range

- Dynamic range is distance in decibels from ENF to amplitude response at maximum
- Does ENF vary with frequency?
 - ☑ Yes

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What should you know?

- **Frequency Distortion**
 - dynamic range or system transfer function
 - not all frequencies are treated equally
 - results in some frequencies being attenuated
- **Transient Distortion**
 - finite vs infinite signal (rise and fall time)
 - not a line spectrum
 - results in spectral components being broaden
- **Amplitude Distortion**
 - system cannot handle the amplitude range
 - peak or amplitude clipping
 - results in harmonics being created

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Intermodulation Distortion

- Driving signal is complex
- Experience nonlinear distortion
- Result called intermodulation distortion

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Intermodulation Distortion

- Suppose input signal has two frequency components
 - ◆ $f_1 = 100$ Hz
 - ◆ $f_2 = 110$ Hz
- Frequency components of output signal include two types of distortion products:
 - ◆ Harmonics
 - ◆ Combination Tones
 - >> Difference tones
 - >> Summation tones

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Intermodulation Distortion

- 1. Harmonics of each frequency component
 - ◆ $1f_1 = 100$ Hz & $1f_2 = 110$ Hz
 - ◆ $2f_1 = 200$ Hz & $2f_2 = 220$ Hz
 - ◆ $3f_1 = 300$ Hz & $3f_2 = 330$ Hz, etc.

Table 7-2. Examples of harmonics and combination tones produced for a complex wave with two frequency components, $f_1 = 100$ Hz and $f_2 = 110$ Hz

HARMONICS OF		COMBINATION TONES	
f_1	f_2	Difference Tones	Summation Tones
$ 1f_1 + 0f_2 = 100$	$ 0f_1 + 1f_2 = 110$	$ 1f_1 - 1f_2 = 10$	$ 1f_1 + 1f_2 = 210$
$ 2f_1 + 0f_2 = 200$	$ 0f_1 + 2f_2 = 220$	$ 2f_1 - 1f_2 = 90$	$ 2f_1 + 1f_2 = 310$
$ 3f_1 + 0f_2 = 300$	$ 0f_1 + 3f_2 = 330$	$ 3f_1 - 1f_2 = 190$	$ 3f_1 + 1f_2 = 410$
etc.	etc.	etc.	etc.

Intermodulation Distortion

- 2. Difference tones:
 - ◆ $|1f_1 - f_2| = 10$ Hz
 - ◆ $|2f_1 - f_2| = 90$ Hz
 - ◆ $|3f_1 - f_2| = 190$ Hz, etc.

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$ 3f_1 + 0f_2 = 300$	$ 0f_1 + 3f_2 = 330$	$ 3f_1 - 1f_2 = 190$	$ 3f_1 + 1f_2 = 410$
etc.	etc.	etc.	etc.



Intermodulation Distortion

● 3. **Summation** tones:

- ◆ $1f_1 + f_2 = 210 \text{ Hz}$
- ◆ $2f_1 + f_2 = 310 \text{ Hz}$
- ◆ $3f_1 + f_2 = 410 \text{ Hz, etc.}$

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etc.	etc.	etc.	etc.

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Combination Tones

● Equation 7.3 defines frequencies of all harmonics, difference tones, and summation tones

- ◆ $mf_1 \pm nf_2$, where
- ◆ m and n are assigned all integer values



Linear Systems

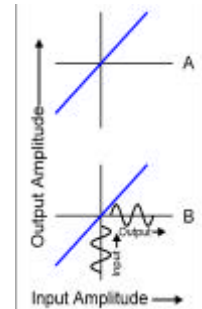
- Linear systems alter only the amplitudes and phases of a signal
- They do produce frequency distortion
- “Frequency response” of an audio device might be described as
 - ◆ 100 Hz to 10,000 Hz
 - ◆ Flat $\pm X \text{ dB}$

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Linear Systems

- Panel A: Input-Output (I/O) function of linear system
 - ◆ As input amplitude increases (e.g., 5 dB), output amplitude increases proportionally (5 dB)
 - ◆ Output amplitude need not equal input amplitude; the change is a proportional one
- Panel B: The characteristics of the input sine wave are preserved faithfully in output sine wave; **Why?**
 - ☑ The system is linear



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